## DEVELOPMENT OF A MOBILE SATELLITE COMMUNICATION UNIT

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#### ABSTRACT

This paper describes a compact (210(W) x 280(H) x 330(D) mm) mobile terminal capable of transmitting voice and data through L-band mobile satellites. The Voice Codec can convert an analog voice to/from digital codes at rate of 9.6, 8 and 4.8 kb/s by an MPC algorithm.

The terminal functions with a single 12 V power supplied by a vehicle battery. The equipment can operate at any L-band frequency allocated for mobile uses in a full duplex mode and will be soon put into a field test via Japan's ETS-V satellite.

#### 1. Introduction

The WARC Mobile 87 has opened a wide path for land mobile satellite communications using L-band frequencies. It is highly expected that satellite systems will take major roles for mobile communications because of their extremely wide coverages [1]. Field data on L-band radio propagation have been accumulated [2] making the mobile satellite services highly realistic. Developments projects for such equipments or systems are going on in various organizations [3,4,5,6,7].

### 2. Basic Design Philosophy

Demands for mobile satellite communications are versatile [6,7,8] and the transmission paths characteristics are often under frequent changes.

Considering above factors, the mobile satellite communication equipments must possess multi-functional capabilities. Such highly adaptive equipments can be most easily implemented by application of DSP (Digital Signal Processing) techniques [8].

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### 3. Basic Design Parameters

### 3.1 Link Budget and Basic Design

Table 3-1 shows link calculations for communications through Japan's ETS-V satellite<sup>[6]</sup> between the Base Earth Station (BES) located at KASHIMA and Mobile Earth Stations (MES) on board automobiles moving around central area of Japan. The calculation is made for the data rate of 9.6 KBPS and for two types of antennas, i.e. omnidirectional (4 dBi gain) and directional (12 dBi gain). The structure and front view of the mobile equipment is each shown in Fig. 3-1 and Photo 3-1.

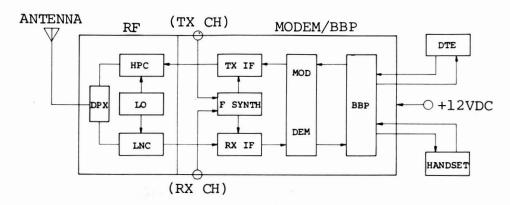


Fig. 3-1 Structure of the Mobile Equipment

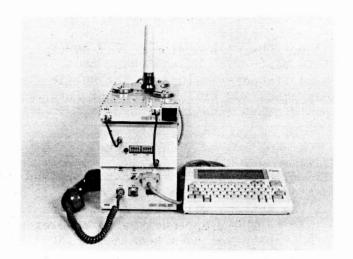


Photo 3-1 External View of the Mobile Equipment

### 3.2 ANTENNA

Two types of antennas are provided for land mobile satellite communications. One is a 3 dBic omni-directional quadrifilar helix type antenna. It can be attached on the vehicle roof easily with magnets and operates without tracking function. Another is a 12 dBic helical antenna for portable communications. It can be mounted on a tri-pod and manually steered for the satellite easily.

Table 3-1 Link Budget

	Omni-directional Dire					
Outside View		1				
Gain	3 dBic	12 dBic				
Polarization	LHCP	LHCP				
Axial Ratio	Less than 2 dB	Less than 2 dB				
Tracking	Unnecessary	Manual				
Radiation Pattern (1625 MHz)	0 dB -5 dB	o dB 5 dB				

## 3.3 RF Unit

# Design concepts

- i) Compactness
  - . The size  $213.5(W) \times 312(D) \times 40(H)$  mm, and the weight 3kg.
  - . A heatsink with a fan keeps the temperature rise within 30°C in a continuous operation.
  - . Compact duplexer (108(W) x 12.5(D) x 20(H) mm) by dielectric resonator technique.
- ii) High performance and flexibility for many experimental uses
  - . The high saturated transmit power and extended linearity achieved with power GaAs FETs (NE345L) usable for carriers of many modulation types.
  - . The low noise GaAs FET (25K569) for the preamplifier stage of the receiver improves the noise figure.
  - . Low phase noise synthesized local oscillators phase locked to a stable reference enable low rate communications.

### Table 3.3 Major RF unit performance

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HPC		
a)	Input frequency	70 MHz band
b)	Input level	O dBm
c)	Output frequency	1626 - 1661 MHz
d)	Output power	20W saturation
	(at DUP output)	
LNC		
a)	Input frequency	1530 - 1559 MHz
b)	Output frequency	70 MHz band

c) Gain

d) Noise figure typical (at DUP input)

DUP Loss transmit side 1.3 dB; receive side 1.4 dB Power requirements

80 dB nominal 2 dB at 25°C

+12 - +15 V DC, 10A (when HPA on)

### High power converter

The transmit BPFs including DUP attenuate more than 100 dB in the 1530 - 1551 MHz receive band to secure 60 dB D/U against local and negligible thermal noise leakages into receive amplifiers from HPC output. HPC output power linearity is shown in Figure 3-3-1.

## Low noise converter

LNC gain and noise figure response is shown in Figure 3-3-2.

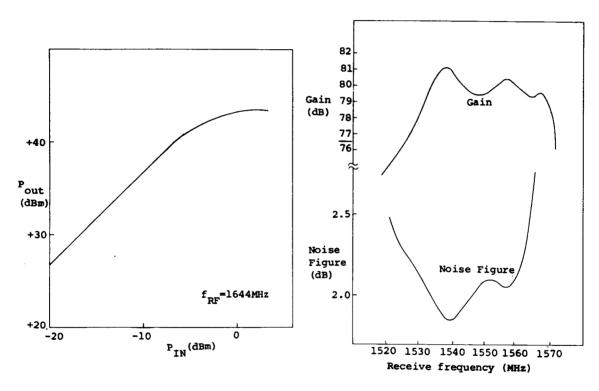


Fig. 3-3-1HPC Output Power Linearity

Fig. 3-3-2LNC GAIN & NF Response

### 3.4 MODEM/BBP

Extensive uses of Digital Signal Processing (DSP) have realized a compact, rate variable MODEM/BBP as depicted in Fig. 3-4. The major design parameters are given in Table 3-4. For the demodulation of receive signals, a coherent detection was adopted because of its good BER performances in such poor C/N conditions as 5 dB E/No. A simple calculation shows that with the local phase noise specified for INMARSAT STANDARD-C system<sup>[7]</sup>, a coherent detection still gives a better BER performance than differential detections in poor C/N conditions for data rates above 4800 BPS.

Table 3-4 MODEM/BBP Parameters

	Modulation Type	QPSK						
	Transmit clock	9900 Hz for 9600 BPS DATA				;		
	Frequency (Hz)	8300 Hz for 8000 BPS DATA						
MOD		5100 Hz for 4800 BPS DATA						
'	TX FILTER Type	Root NYQUIST		TX IF	DSP MODEM	BBP		
1	TX IF FREQ (MHz)	70 + 15 (5 KHz step)					,	
1	TX IF Level	010 dBm/75	ТX	IF FREQ	<del>     -    -                            </del>	TX BBP		VOICE
	RX IF FREQ (MHz)	70 + 15 MHz (5 KHz step)		CONTV D/A	MOD	IX BB1		CODEC
	RX IF Level	-50 dBm/Ch						
DEM	RX FILTER Type	Root NYQUIST					111	
1	Detection	Coherent Detection	mv	C" -		+ - +		
	Method of Word	By Frame Synch.	<u>TX</u>		CLK PH	CLK FEC	111	
i	Synchronization		RX	CH F.SYNTH	RX CLK ERR	CODEC		
1	FEC	R=1/2 Convolutional Coding,	(		RECOVERY	GEN (LSI)		
BBP	l.	Constraint Length; 4 bits,	-	4f CLI	N TT	<del>                                     </del>		
1	ł	3 bit soft decision, Viterbi		RX IF		<u> </u>	_	
		Decoding, Coding Gain 3.5 dB		- PRO		RX RX	[ ] [ [ [	
1		at BER 10-4	RX :	11 A/D	CH	FR BBF		DTE
				CONV	- FIL	+- sime_	<u>-                                    </u>	DIE
				Note:	Symbol st	ands for a DSP (	(uPD77P20	)
				No cc,		v NEC)	• •	

Fig. 3-4 Structure of MODEM/BBP

### 3.5 VOICE CODEC

A 4.8 to 16 Kb/s high quality voice codec is implemented on NEC's single chip advanced floating point digital signal processor (ASP) UPD77230[9]. The codec is a multi-pulse exited linear predictive coder (MPELPC) based on a pulse search method which integrates the conventional cross correlation method [10] and the spectrum peak emphasis (SPE) method [11]. By adopting the SPE method, the segmental signal to noise ratio (SNRSEG)[12] has been improved by approximately 2.2 dB to 2.8 dB for bit rates below 7.2 Kb/s. The total coding delay is only 45 msec. This new codec can achieve speech coding at any rate between 16 and 4.8 Kb/s. Details of the codec specifications are shown in Table 3-5.

The codec is mounted on a single board as shown in Photo 3-5 and the structure illustrated in Fig. 3-5-1. The codec consists of a single ASP for digital signal processing, a single chip —- LAW PCM codec for analogue interface, clock and timing circuits, and a digital line interface. The dimension of the board is 125 mm by 145 mm.

Fig. 3-5-2 shows the two  ${\rm SNR}_{\rm SEG}$  scores tested for eighteen short English sentences spoken by three American males and twelve short Japanese sentences spoken by two female and one male Japanese.

Table 3-5 Codec Specifications

NO	ITEM	SPECIFICATION
1	Sampling	8kHz (16K/9.6K/8K) 6.4kHz (7.2K/4.8K)
2	Quantization	8bits (255)
3	Frame Period	20 msec
4	LPC Analysis	Auto correlation method
5	LPC Order	12th (16K/9.6K/8K) 9th (7.2K/4.8K)

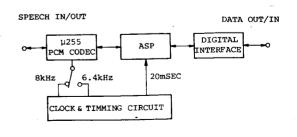
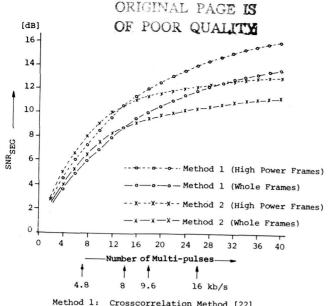


Fig. 3-5-1 The Block Diagram of Coder Hardware



Method 1: Crosscorrelation Method [22]
Method 2: SPE Method [23]

Fig. 3-5-2 Results of SNR<sub>SEG</sub>

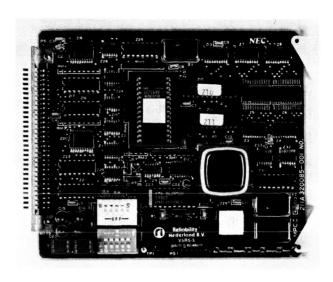


Photo 3-5 View of the New Codec Board

## Conclusion

A compact multifunctional mobile terminal has been developed for field tests with Japan's ETS-V satellite.

### REFERENCES

- [1] Bullock, C. 1986. The inviolable market? Mobile communications need satellites. In SPACE MARKETS Winter/No. 4 Quarterly.
- [2] R.W. Huck, J.S. Butterworth and E.E. MATT, 1983. "Propagation Measurements for Land Mobile Satellite Services" IEEE VTC
- [3] T. Ikegami, R. Suzuki, Y. Hase, S. Ohmori and K. Kosaka, 1987
  "Land Mobile Communication Experiments with ETS-V satellite"
  IEEE VTC
- [4] T. SAKAI, H. KOMAGATA, N. TERADA, and E. HAGIWARA, 1987.
  "Compact Mobile Terminals for Multi-Beam Mobile Satellite
  Communication Systems" IEEE GLOBECOM, TOKYO
- [5] Y. YASUDA, M. OHASHI, F. SUGAYA, H. OKINAKA, Y. HIRATA, 1987. "Field Experiment on Digital Maritime Satellite Communication Systems" IEEE GLOBECOM TOKYO
- [6] S. MIURA, M. WAKAO, N. FUTAGAWA, 1985. "L-BAND Transponder of ETS-V for Experimental Mobile Satellite System" IEEE ICC
- [7] INMARSAT STANDARD-C Specification
- [8] J.H. LODGE, D. BOUDREAU, 1986. "The Implementation and Performance of Narrowband Modulation Techniques for Mobile Satellite Applications" IEEE ICC
- [9] T. Nishitani, 1986. "Advanced Single-Chip Signal Processor", Proc. ICASSP '86, Apr. 1986
- [10] T. Araseki, 1983. "Multi-Pulse Excited Speech Coder Based on Maximum Crosscorrelation Search Algorithm", GLOBECOM 83, Dec. 1983.
- [11] T. Taguchi, 1987. "Implementation of a 4.8 16 Kbps Multi-Pulse Codec on a Single Chip DSP Based on Spectrum Peak Emphasis", 30th Midwest Symposium on Circuits & Systems
- [12] T. Watanabe, 1982. "Comparison of Performance of Voiceband Codecs by Several Speech Quality Measures", Trans. of the committee on Speech Research, The Acoustical Society of Japan.

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